A2.03 AdTran NetVanta 1355/6355 SIP/TDM/IP Appliance



1 Important Notes

- Check the SIP 3rd Party Validation Website for current validation status. The SIP 3rd Part Validation Website can be viewed at:
- http://testlab.vonexus.com
- It is recommended to make basic changes to the Interactive Intelligence supplied .cfg file then utilize the command line interface for further configuration. Not all options are exposed using the web interface.
- As this unit is a critical piece of the SIP infrastructure, it is highly recommended that DHCP not be used. A static IP address is the preferred method and is how the unit was configured during validation.
- While there are TCP options in the command line, they are not recommended for use by AdTran, and were not tested.
- T.38 is supported for 4 concurrent faxes on the Digital span *only*. T.38 is not supported for the FXO spans.
- When using managed phones the name of the phone in IA must match the
 extension of the phone. This is required because the phone will use the name
 in its registration message, and in a standalone failover scenario, the
 registration name is how the AdTran device will be able to contact the other
 internal stations. If something else is used, internal stations would not be
 able to dial other internal stations, because the number dialed would not
 match the registration entries.

2 Vendor Documentation

Documentation can be found on the CD shipped with the AdTran NetVanta 1355/6355, or on the AdTran website http://www.adtran.com.

3 Validated Firmware Version

A2.03.00.SC.E

4 Install

Download the AdTran NetVanta 1355/6355 files from the Interactive Intelligence Testlab website for the AdTran:

http://testlab.inin.com/compatibilityfiles external/production/documents/AdTranNetVanta.zip

Contained in the zip file will be the validated version of firmware (.biz), as well as any supplemental configuration files, and a set of sample (.cfg) files.

5 Configuration

Methods:

- Manipulation of the supplied .cfg file, then uploading it via the web interface. This method
 is recommended for a quick start scenario. The Interactive Intelligence supplied .cfg file,
 with just a few modifications will get a basic setup up and running. Please note, the
 supplied .cfg file is not meant to be a drop in configuration for every environment, it is
 provided as reference only.
- Web interface. (This method was not used to configure the unit during validation) There
 are many advanced options that are not exposed in the web interface. Caution should be
 exercised and the AdTran documentation should always be referenced when using the
 web interface configuration option.
- TFTP. This has not been tested.

Initial Setup:

- Unzip the ZIP file containing the AdTran NetVanta 1355/6355 configuration files and firmware.
- Follow the AdTran instructions (on the CD or in the booklet shipped with the NetVanta or
 in) for getting an IP address assigned to the device.
- Modify the I3 provided .cfg file as appropriate.
- Start a web browser and type in the IP address of the NetVanta.
- The default user name is "admin" and the password is "password". It is recommended to change the password when possible for security reasons.

Upload Current Firmware (biz file):

- From the main web screen, select the *Utilities* breakout tab from the icons on the left side of the page.
- Select Firmware from the expanded options under Utilities.
- Where it reads *Select firmware file:* click the browse button, navigate to the appropriate .biz file, click ok, then click the *Upload* button.
- After the validated firmware has been applied the NetVanta will require a restart. Once
 the restart is complete, return to the web interface. The opening screen should show the
 uploaded firmware version.

Upload Sample Config File:

- From the main web screen, select the *Utilities* breakout tab from the icons on the left side of the page.
- Select Configuration from the expanded options under Utilities.
- Where it reads *Upload Config:* click the browse button, navigate to the appropriate .cfg file, click ok, then click the *Upload* button.
- After the configuration has been applied the NetVanta will require a restart.
- Please note that this will put in the configuration from the provided files. Make sure to give it the appropriate IP addresses for the network in question, or the device may get an invalid address.
- This is the only configuration that will be done via the web interface, due to config options that are not exposed except via the command line.

Changing the Configuration:

This section will go through the sample ISDN NI2 PRI User side configuration for managed phone support, and highlight the significant configuration options. All other options can be found in the AdTran User Documentation.

```
! Boot ROM version 14.02.00
! Platform: NetVanta 6355, part number 1200740E1
! Serial number H15F6449
hostname "NV1355"
enable password encrypted 28207f837283dac7c60316194a65e4b3d52a
clock timezone -5-Eastern-Time
ip subnet-zero
ip classless
ip routing
ip domain-name "domain.com"
This should be the domain name relative to the business
ip domain-proxy
ip name-server 192.168.1.230 192.168.1.231
This should be the DNS server
no auto-config
event-history on
no logging forwarding
no logging email
service password-encryption
username "admin" password encrypted "2720aa4ceca5b57915c1839aa58a7da49d2f"
username "polycomftp" password encrypted
"4048e1bf5bde64becb908f2203161619f745"
banner motd #
     Important
Web username/password is configured to admin/password.
Enable and Telnet passwords are configured to "password".
Please change them immediately.
The VLAN 1 interface is enabled with an address of 10.10.10.1
Telnet/SSH access is also enabled.
!
ip firewall
no ip firewall alg msn
no ip firewall alg h323
no dot11ap access-point-control
vlan 1
 name "Default"
interface eth 0/0
```

```
description Uplink To Outside
 no ip address
 no shutdown
 no lldp send-and-receive
interface eth 0/1
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/2
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/3
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/4
 no shutdown
 switchport mode trunk
interface eth 0/5
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/6
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/7
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/8
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/9
 spanning-tree edgeport
```

```
no shutdown
 switchport mode trunk
interface eth 0/10
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/11
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/12
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/13
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/14
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/15
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/16
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/17
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/18
 no shutdown
 switchport mode trunk
```

```
interface eth 0/19
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/20
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/21
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/22
 spanning-tree edgeport
 no shutdown
 switchport mode trunk
interface eth 0/23
 no shutdown
 switchport mode trunk
interface eth 0/24
 no shutdown
 switchport mode trunk
interface gigabit-eth 0/1
 no shutdown
 switchport mode trunk
interface gigabit-eth 0/2
 no shutdown
 switchport mode trunk
interface vlan 1
 ip address 192.168.1.5 255.255.0.0
This ends up being the IP of the device... please note that this is only using
one vlan for testing purposes, business needs may require alternatives
(voice vlan, etc...)
 media-gateway ip primary
```

```
no shutdown
interface t1 1/1
 description NI2 PRI
This is the type of ISDN protocol
 system-timing primary
This has the system clocking being driven by the line
 tdm-group 1 timeslots 1-24 speed 64
This allots all ISDN channels to use with this port
 no shutdown
interface t1 2/1
 shutdown
interface pri 1
 description pri 1
This is the description for the PRI interface (which is a logical interface
which is linked to the interface t1 1/1 above)
 isdn name-delivery setup
 connect t1 1/1 tdm-group 1
This is the logic that links the PRI section with the T1 section)
 role user
This tells the line to be user side (vs. network side)
 no shutdown
interface fxs 0/1
 description "Phone Station"
 no shutdown
interface fxs 0/2
 no shutdown
interface fxo 0/1
 description "715-4223"
 no shutdown
interface fxo 0/2
 description "test fxo2"
 no shutdown
isdn-group 1
isdn-group 2
 connect pri 1
ip route 0.0.0.0 0.0.0.0 192.168.0.1
This is a standard default route
```

ip route 192.168.1.250 255.255.255.255 null 0

This route is created specifically for the managed phone option, it will allow the phones to register to a route that is known dead, but required for the device to respond

no ip tftp server
no ip tftp server overwrite
ip http server
ip http secure-server
no ip snmp agent
no ip ftp server
ip ftp server default-filesystem flash
no ip scp server
no ip sntp server
!
!
!
!
!

voice feature-mode network

This needs to be set to network to allow the inbound to FXS station in survivability mode

voice forward-mode network

Ţ

This needs to be set to network to allow the inbound to FXS station in survivability mode

```
conference
 dnd
 external-fwd
 forward
 hold
 hotel
 logout-group
 message-waiting
 overhead-paging
 redial
 remote-fwd
 return-last-call
 system-speed
 transfer
voice codec-list "7xx Options"
 default
 codec g729
 codec q711ulaw
 codec g711alaw
voice trunk T01 type sip
 description "SIP TRUNK"
 caller-id-override number-inbound 9 if-no-cpn
 sip-server primary 192.168.1.10
This sets the primary sip server for incoming calls to be attempted
 sip-server secondary 192.168.1.11
This sets the secondary sip server for incoming calls to be attempted
This is always done as a primary/secondary option, there is no "parking"
method
 authentication username "test" password encrypted
"1f1b093b2b136f39041ac568f6e7eec6ada1"
 register range 8001 8002
 check-supported replaces
 codec-group "7xx Options"
Associate a codec group with the line
 default-ring-cadence internal
voice trunk T02 type analog supervision loop-start
 description "FXO"
 caller-id
 caller-id-override number-inbound 1231234 if-no-cpn
 trunk-number 7154223
The number associated with the analog line
 connect fxo 0/1
 modem-passthrough
 rtp delay-mode adaptive
 codec-group "7xx Options"
voice trunk T03 type isdn
 description "NI2 PRI"
 resource-selection circular descending
Order of lines selected for outgoing PRI calls
```

```
connect isdn-group 2
 modem-passthrough
 t38
Allows T.38 faxing
 rtp delay-mode adaptive
 codec-group "7xx Options"
voice trunk T04 type analog supervision loop-start
 description "FXO2"
 caller-id
 caller-id-override number-inbound 1231234 if-no-cpn
 trunk-number 7154222
 connect fxo 0/2
 modem-passthrough
 rtp delay-mode adaptive
 codec-group "7xx Options"
voice grouped-trunk "SIP GROUP"
This section associates the SIP trunk with its routing and number options
 no description
 trunk T01
Shows which trunk to associate with the group
 accept 6739 cost 0
Will accept this number (6739) bound to trunk at route weight cost of 0
(best), should be set to PRI number
 accept $ cost 0
Will accept any number bound to SIP trunk T01 at weight cost 0, forces all
calls to go out SIP trunk first (so station calls go to IC server first)
 reject 976-XXXX
 reject 1-900-NXX-XXXX
 reject 1-976-NXX-XXXX
Will not allow any 900 or 976 numbers to be dialed or accepted
Ţ
voice grouped-trunk FXO
 description "FXO Ports"
 trunk T02
 trunk T04
 accept 911 cost 0
This forces 911 calls to go out the FXO, rather than the PRI, for
emergencies, this may be adjusted according to business needs
 accept $ cost 400
This assures that calls can use the FXO port to go outbound, but will
 accept 7154222 cost 0
This routs calls to this number out the FXO port specifically, this may or may
not fit the business plan, but can be useful in some circumstances (e.g. fax
line)
 reject 976-XXXX
 reject 1-900-NXX-XXXX
 reject 1-976-NXX-XXXX
 reject 6739
This will cause any calls bound for the PRI not to go through this FXO group
Ţ
```

```
voice grouped-trunk "NI2 PRI"
 no description
 trunk T03
 accept $ cost 100
This causes all calls bound for the PRI to be accepted with weight cost of
100, this sits between the SIP and FXO values to attempt to force all calls to
go through IC first (but any sent from IC will go out this trunk because they
came in on the SIP one)
 reject 976-XXXX
voice user 8001
 connect fxs 0/1
 first-name "test"
 last-name "fxs"
 password encrypted "2226c76989b7cc49da5ab592196a551b8c23"
 no special-ring-cadences
 coverage external 2222222
 codec-group "7xx Options"
This is the setup for a user via the AdTran device (in this case an FXS user)
with associated options (this can also be done with an IP phone)
voice user 8002
 connect fxs 0/2
 first-name "Fax"
 last-name "Machine"
 password encrypted "2226c76989b7cc49da5ab592196a551b8c23"
 no special-ring-cadences
 modem-passthrough
 t38
 t38 error-correction redundancy
 codec-group "7xx Options"
voice ring-group 1234
This defines the survivability inbound calling direction
 description backup for incoming calls if no IC server
 num-rings 4
 member 8001
 login-member 8001
This is the user to which the inbound PRI calls should go in survivability
mode
 alias 6739
This is the incoming number (PRI number) that this section will look for in
survivability mode
ip sip
no ip sip registrar authenticate
This prevents the device from forcing authentication to be added to the
```

```
standalone survivability)
NOTE: This turns off the device authentication, so any compatible device
that directly points to it can make outbound calls
ip sip proxy
ip sip proxy domain "qfun.com"
ip sip proxy sip-server primary 192.168.1.250 tcp
This is the proxy server address for the device, which was set to the null
route above
ip sip proxy emergency-call-routing accept $
This allows calls to be made in survivability mode
ip sip proxy failover accept-registrations
This allows the device to accept phone registrations for survivability mode
no ip sip grammar supported 100rel
ip rtp symmetric-filter
This allows failover with the FXS stations to the secondary SIP proxy
ip rtp quality-monitoring
ip rtp quality-monitoring sip
line con 0
 no login
line telnet 0 4
 login
 password encrypted 171f5669b452a522d8189e510076754304ae
 line-timeout 0
 no shutdown
line ssh 0 4
 login local-userlist
 no shutdown
sntp server 192.168.1.2 version 3
Ţ
end
```

proxy list (this will allow managed phones to register to the device for

There are also a few variants from the above configuration for certain other options.

For ISDN network side, replace the interface section for "pri 1" with the following.

```
interface pri 1
 description pri 1
 isdn name-delivery setup
 connect t1 1/1 tdm-group 1
 role network b-channel-restarts disable
 no shutdown
For T1 CAS, replace the interface section for "t1 x/x" with the following
interface t1 1/1
 description NI2 PRI
 clock source internal
 system-timing primary
 tdm-group 1 timeslots 1 speed 64
 tdm-group 2 timeslots 2 speed 64
 tdm-group 3 timeslots 3 speed 64
 tdm-group 4 timeslots 4 speed 64
 tdm-group 5 timeslots 5 speed 64
 tdm-group 6 timeslots 6 speed 64
 tdm-group 7 timeslots 7 speed 64
 tdm-group 8 timeslots 8 speed 64
 tdm-group 9 timeslots 9 speed 64
 tdm-group 10 timeslots 10 speed 64
 tdm-group 11 timeslots 11 speed 64
 tdm-group 12 timeslots 12 speed 64
 tdm-group 13 timeslots 13 speed 64
 tdm-group 14 timeslots 14 speed 64
 tdm-group 15 timeslots 15 speed 64
 tdm-group 16 timeslots 16 speed 64
 tdm-group 17 timeslots 17 speed 64
 tdm-group 18 timeslots 18 speed 64
 tdm-group 19 timeslots 19 speed 64
 tdm-group 20 timeslots 20 speed 64
 tdm-group 21 timeslots 21 speed 64
 tdm-group 22 timeslots 22 speed 64
 tdm-group 23 timeslots 23 speed 64
 tdm-group 24 timeslots 24 speed 64
 no shutdown
and replace the PRI voice trunk with the following
voice trunk T04 type t1-rbs supervision wink role network
 description "CAS"
 connect t1 1/1 tdm-group 1
 connect t1 1/1 tdm-group 2
 connect t1 1/1 tdm-group 3
 connect t1 1/1 tdm-group 4
 connect t1 1/1 tdm-group 5
 connect t1 1/1 tdm-group 6
 connect t1 1/1 tdm-group 7
 connect t1 1/1 tdm-group 8
 connect t1 1/1 tdm-group 9
 connect t1 1/1 tdm-group 10
```

```
connect t1 1/1 tdm-group 11 connect t1 1/1 tdm-group 12 connect t1 1/1 tdm-group 13 connect t1 1/1 tdm-group 14 connect t1 1/1 tdm-group 15 connect t1 1/1 tdm-group 16 connect t1 1/1 tdm-group 17 connect t1 1/1 tdm-group 18 connect t1 1/1 tdm-group 19 connect t1 1/1 tdm-group 20 connect t1 1/1 tdm-group 21 connect t1 1/1 tdm-group 22 connect t1 1/1 tdm-group 23 connect t1 1/1 tdm-group 24 rtp delay-mode adaptive
```

In each case, please verify that the voice grouped trunks contain the proper references for the new interface identifiers.

Samples for NI2 user, NI2 network, and T1 CAS will be provided in the firmware files zip archive. Other ISDN protocols are available, just not as a sample. Check the AdTran documentation for details.

It is also possible when not using managed phones to manually assign a user to each phone on the device for standalone failover options. This will allow the failover to happen more efficiently, but will require an entry both on the IC server, and the AdTran device for *every* phone being used.

6 Putback Transfer

Putback transfer is not supported by AdTran for this device at this time.

7 Security

TLS/SRTP is not supported by AdTran for this device at this time.